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NETWORK ALTERNATE
ROUTING SYSTEM

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Takeo ORUI

Enclosed are:

- [X] Specification, Claim(s), and Abstract (32 pages).
- [X] Formal drawings (3 sheets, Figures 1-3).
- [X] Declaration and Power of Attorney (2 pages).
- [X] Assignment of the invention to NEC CORPORATION.
- [X] Assignment Recordation Cover Sheet.
- [X] Claim for Convention Priority w/1 certified document.

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Internet Protocol Network Alternate Routing System

BACKGROUND OF THE INVENTION

1. Field of the Invention:

5 The present invention relates to an Internet protocol network alternate routing system for rerouting voice signals that are transmitted from an extension telephone from a signal path connected to an Internet protocol network to a leased line or an office line.

10 2. Description of the Related Art:

 In recent years, Internet protocol (herein-below abbreviated as IP) networks, in which data and voice signals are integrated and transmitted, have been used in business networks, and efforts are now being made to
15 reduce the operating costs of these networks.

 In a conventional system that uses an IP network, networks are connected together via analog interfaces or digital interfaces by means of gateway devices, and these gateway devices are further connected to private branch
20 exchanges (herein-below abbreviated PBX) that are connected to telephones and computer terminals. Telephones and computer terminals are therefore connected to each other by way of IP networks.

 In a PBX that is provided within an IP network,
25 voice signals are converted to packets and then transmitted. However, there has been the problem that the

real-time reproduction of packet voice signals is prevented by the occurrence of trouble or congestion within the IP network, and the level of service is therefore reduced.

5 This problem has occurred because voice signals could not be rerouted from a signal path that passes by way of an IP network to a TDM (Time Division Multiplexer) leased line or a public telecommunication network. This type of alternate routing could not be realized in a
10 conventional IP network.

SUMMARY OF THE INVENTION

 The present invention was developed in view of the above-described problem of the prior art and has the
15 object of providing an IP network alternate routing system that is capable of rerouting voice signals that are transmitted from an extension telephone to a TDM leased line or to a public telecommunication network when trouble or congestion occurs in an IP network.

20 To achieve the above-described object, the present invention features:
 an Internet protocol network alternate routing system having an extension telephone, a plurality of networks provided with one Internet protocol network for
25 transmitting voice signals from the extension telephone, and an exchange for controlling connections between the

extension telephone and the plurality of networks,
the exchange, upon detecting a state of congestion of the
Internet protocol network, automatically switches the
network that is connected with the extension telephones
5 to a network other than the Internet protocol network.

The exchange is provided with:

a plurality of signal paths for connecting the plurality
of networks with the extension telephone;

an alternate routing control unit for determining the
10 network that is to be connected with the extension
telephone;

a call control unit for establishing a link between the
network that has been determined by the alternate routing
control unit and the extension telephone;

15 a traffic control unit for detecting the state of
congestion of a network for which a link was established
by the call control unit; and

a switch control unit that controls the connection
between the extension telephone and a signal path among
20 the plurality of signal paths that is connected to a
network, based on the detection results in the traffic
control unit.

The exchange further includes:

a voice converter that, when a link has been established
25 with a partner node by the call control unit and voice
signals are transmitted from the extension telephone by

way of the switch control unit, converts the transmitted voice signals to packets, assigns port numbers to each of the packet voice signals, and outputs the voice signals; wherein the traffic control unit adds Internet protocol addresses that are outputted from the call control unit to voice signals outputted from the voice converter to generate and output Internet protocol packets; and further, the exchange includes:
an RTP unit for adding RTP headers to Internet protocol packets outputted from the traffic control unit and outputting the result; and
a network driver for transmitting on an Internet protocol network Internet protocol packets to which RTP headers have been added and that have been outputted from the RTP unit.

In addition, in a case in which Internet protocol packets to which RTP headers have been added are transmitted from the Internet protocol network, the RTP unit removes the RTP headers and outputs to the traffic control unit;
the traffic control unit converts the Internet protocol addresses contained in the Internet protocol packets that are outputted from the RTP unit to port numbers of ports in the voice converter, and outputs each of the Internet protocol packets that are assigned by port in the voice converter to a respective port in the voice converter;

and the voice converter converts the Internet protocol packets that are outputted from the traffic control unit to voice signals, reconfigures the converted voice signals, and transmits to the extension telephone by way
5 of the switch control unit.

In addition, the traffic control unit detects a packet loss rate that is contained in a sender report packet that is transmitted from the Internet protocol network, and informs the call control unit that the
10 packet loss rate has exceeded a predetermined set value if the packet loss rate exceeds the set value; and the call control unit is provided with a counter in which the count value is incremented with each notification from the traffic control unit that the packet loss rate
15 has exceeded a set value, and the call control unit does not establish a link between the Internet protocol network and the extension telephone if the count value exceeds a predetermined set value.

In addition, the packet loss rate in the sender
20 report packet is variable.

The switch control unit may also be manually switched.

[The invention is further features] an announcement trunk for reporting switching of the signal path to the
25 extension telephone when the network that was determined by the alternate routing control unit is in a congested

state.

(Operation)

In the present invention configured according to the above description, when voice signals transmitted from the extension telephone are transmitted on an Internet protocol network, the state of congestion of the Internet protocol network is judged by the call control unit based on sender packets that are transmitted from the Internet protocol network; and when the Internet protocol network is judged to be in a congested state, the signal path of the voice signals is switched by the switch control unit and the voice signals are rerouted from the Internet protocol network to a network other than the Internet protocol network.

Voice signals that are transmitted from the extension telephone are thus automatically rerouted to a network other than the Internet protocol network when the Internet protocol network is in a congested state, and the level of service of the system can therefore be maintained.

The above and other objects, features, and advantages of the present invention will become apparent from the following description based on the accompanying drawings which illustrate examples of preferred embodiments of the present invention.

BRIEF DESCRIPTION OF THE DRAWINGS

Fig. 1 shows one embodiment of the Internet protocol network alternate routing system of the present invention;

5 Fig. 2 shows one configuration of the RTCP sender report packet that is transmitted from a LAN upon establishing a link in the call control unit shown in Fig. 1; and

Fig. 3 shows another embodiment of the Internet protocol network alternate routing system of the present invention.

DETAILED DESCRIPTION OF THE PREFERRED EMBODIMENTS

An explanation of embodiments of the present invention is next presented with reference to the accompanying figures.

Fig. 1 shows one embodiment of the IP network alternate routing system of the present invention. As shown in Fig. 1, the present embodiment is made up by:
20 an extension telephone 1 that transmits PCM (Pulse Code Modulation) voice signals; LAN (Local Area Network) 10, which is an IP network for transmitting the PCM voice signals that are transmitted from extension telephone 1; leased line network 13, which is a line different from
25 LAN 10, for transmitting PCM voice signals that are transmitted from extension telephone 1; leased line trunk

12 for rerouting PCM voice signals that are transmitted
from extension telephone 1 to leased line network 13; and
PBX (Private Branch Exchange) 11, which is an exchange
for connecting the signal path of PCM voice signals that
5 are transmitted from extension telephone 1 to one of LAN
10 and leased line network 13.

In addition, extension telephone 1 is assigned in
advance with a service class for determining the signal
path by which PCM voice signals are to be transmitted.

10 PBX 11 is made up of: central processing unit 3 that
designates the signal path by which PCM voice signals are
transmitted from extension telephone 1; switch control
unit 2 for switching the signal path that is path-
connected to extension telephone 1 based on the signal
15 path designated by central processing unit 3; LAN driver
9, which is a network driver at the interface for LAN 10;
alternate routing control unit 4 for determining the
signal path of the PCM voice signals that are transmitted
from extension telephone 1 in accordance with the service
20 class that is assigned in advance to extension telephone
1 and notifying the determined signal path to central
processing unit 3; call control unit 5 for both
converting the number of the partner node that was dialed
by extension telephone 1 to an IP address and outputting
25 the result when the signal path notified to central
processing unit 3 is signal path A to LAN 10 and

establishing a link to the partner node by way of LAN driver 9 and LAN 10; voice converter 6 for converting to packets the PCM voice signals that are transmitted from extension telephone 1 by way of switch control unit 2

5 when a link to the partner node has been established by call control unit 5, assigning port numbers to each of the PCM voice signal packets, and outputting the PCM voice signals; traffic control unit 7 for adding IP addresses outputted from call control unit 5 to PCM voice

10 signals that are outputted from voice converter 6 to generate and output IP packets; RTP unit 8 for adding RTP (Real-time Transport Protocol) headers to IP packets outputted from traffic control unit 7 and transmitting to the partner node by way of LAN driver 9 and LAN 10. When

15 IP packets to which RTP headers have been added are transmitted from a partner node to LAN driver 9 by way of LAN 10, the RTP headers that have been added to the IP packets are removed by RTP unit 8, the IP headers included in the IP packets are converted by traffic

20 control unit 7 to port numbers of voice converter 6, the IP packets that are assigned by port number in voice converter 6 are outputted to each port in voice converter 6, the IP packets assigned by port number are converted to PCM voice signals by voice converter 6, and the

25 converted PCM voice signals are reconfigured and transmitted by way of switch control unit 2 to extension

telephone 1.

The signal paths that are path-connected to extension telephone 1 by switch control unit 2 are constituted by signal path A, over which PCM voice signals are transmitted by way of LAN 10, and signal path B, over which PCM voice signals are transmitted by way of leased line network 13.

In switch control unit 2, signal paths may be switched either automatically in accordance with instructions from central processing unit 3 or by manual operation.

In addition, the RTP headers that are added to IP packets by RTP (Real-time Transport Protocol) unit 8 are provided for realizing real-time transmission of IP packets by RTP on LAN 10.

When IP packets are transmitted by RTP, RTCP (RTP Control Protocol) for confirming transmission or monitoring IP packets is appended when transmitting IP packets or when notifying of the establishment of a link to a partner node. As a result, an RTCP sender report packet, which is a sender report packet, is transmitted from LAN 10 to LAN driver 9 when transmitting IP packets or when reporting the establishment of a link to a partner node, and in call control unit 5, a link is established to the partner node by way of LAN 10 based on this RTCP sender report packet.

Explanation is next presented regarding signal transmission operations in the IP network alternate routing system that is configured according to the above-described description.

5 Explanation is first presented regarding the operations when transmitting PCM voice signals from extension telephone 1 to a partner node.

When the number of a partner node is dialed by extension telephone 1, an inquiry of the order of
10 alternate routing is sent from central processing unit 3 to alternate routing control unit 4.

Based on the service class that was assigned in advance to extension telephone 1, the first signal path is then determined in alternate routing control unit 4 to
15 be signal path A, which is connected to the partner node by way of LAN 10, which is path-connected to voice converter 6. Central processing unit 3 is then notified that the first signal path is signal path A.

In some cases, the first signal path is determined,
20 according to the service class that is assigned in advance to extension telephone 1, to be signal path B, which is connected to the partner node by way of leased line network 13 and path-connected to leased line trunk
12.

25 Central processing unit 3 instructs call control unit 5 to establish a link to the partner node by way of

LAN 10.

Upon being instructed by central processing unit 3 to establish a link, call control unit 5 converts the number of the partner node that was dialed by extension
5 telephone 1 to an IP address and establishes a link via LAN driver 9 and LAN 10 to the partner node having the converted IP address.

At this time, the IP address that was converted from the number of the partner node by call control unit
10 5 is outputted from call control unit 5 to traffic control unit 7.

Upon notification from call control unit 5 that a link has been established to the partner node by way of LAN driver 9 and LAN 10, an RTCP sender report packet
15 that is defined by RTCP is transmitted from LAN 10 to LAN driver 9, following which the RTCP sender report packet that was transmitted to LAN driver 9 is transmitted from LAN driver 9 to traffic control unit 7 by way of RTP unit 8.

20 In traffic control unit 7, the state of congestion of LAN 10 is checked based on the RTCP sender report packet that was transmitted from LAN 10, and the state of congestion of LAN 10 that was checked is then reported to call control unit 5.

25 In call control unit 5, the state of congestion of LAN 10 is judged based on the check of the RTCP sender

report packet at traffic control unit 7.

If it is hereupon judged in call control unit 5 that LAN 10 is not in a congested state and that a link has been established to the partner node by way of LAN 10, the establishment of a link is reported from call control unit 5 to central processing unit 3.

Central processing unit 3 next instructs switch control unit 2 and call control unit 5 to reproduce the PCM voice signals that were transmitted from extension telephone 1, following which extension telephone 1 and voice converter 6 are path-connected by switch control unit 2, and PCM voice signals are subsequently transmitted from extension telephone 1 to the partner node by way of LAN 10.

When the reproduction of PCM voice signals that were transmitted from extension telephone 1 is instructed from central processing unit 3, the PCM voice signals transmitted from extension telephone 1 are converted to packets in voice converter 6, a port number is assigned to each of the packet PCM voice signals, and the packet PCM voice signals that have been assigned port numbers are outputted to traffic control unit 7.

Traffic control unit 7 next adds IP addresses, which have been outputted from call control unit 5, to the PCM voice signals, which have been outputted from voice converter 6, to generate IP packets; and outputs

the generated IP packets to RTP unit 8.

In RTP unit 8, RTP headers are then added to the IP packets that have been outputted from traffic control unit 7, and the IP packets to which RTP headers have been
5 added are then transmitted to the partner node by way of LAN driver 9 and LAN 10.

If, on the other hand, call control unit 5 judges that LAN 10 is in a congested state and a link is not established to the partner node by way of LAN 10, central
10 processing unit 3 is notified from call control unit 5 that a link has not been established.

In the event of faults in the interface with LAN 10 in the partner node as well, a link is not established with the partner node, and call control unit 5 therefore
15 similarly notifies central processing unit 3 that a link has not been established.

Upon notification from call control unit 5 to central processing unit 3 that a link has not been established on LAN 10 with the partner node, central
20 processing unit 3 sends an inquiry to alternate routing control unit 4 for a second signal path.

Based on the service class that was assigned in advance to extension telephone 1, alternate routing control unit 4 next determines the second signal path to
25 be signal path B, which is connected to the partner node by way of the leased line network 13 that is path-

connected to leased line trunk 12, and notifies central processing unit 3 that the second signal path is signal path B.

Central processing unit 3 next instructs switch
5 control unit 2 to reroute PCM voice signals that are transmitted from extension telephone 1 to signal path B, following which extension telephone 1 and leased line trunk 12 are path-connected by switch control unit 2 and PCM voice signals are subsequently transmitted from
10 extension telephone 1 to the partner node by way of leased line network 13 that has been rerouted to signal path B.

Explanation is next presented regarding operation when IP packets to which RTP headers are added are
15 transmitted from the partner node to LAN driver 9 via LAN 10.

When IP packets to which RTP headers have been added are transmitted to LAN driver 9 by way of LAN 10, the RTP headers that were added to the IP packets that
20 are transmitted to LAN driver 9 are removed in RTP unit 8, following which the IP packets from which the RTP headers have been removed are outputted to traffic control unit 7. In traffic control unit 7, the IP headers that are included in the IP packets that have been outputted from
25 RTP unit 8 are next converted to port numbers of voice converter 6, and each of the IP packets that have been

assigned to a port number of voice converter 6 are then outputted to the corresponding port provided in voice converter 6.

In voice converter 6, the IP packets outputted from traffic control unit 7 are converted to PCM voice signals at each of the incorporated ports, whereby the converted PCM voice signals are reconfigured and transmitted to extension telephone 1.

The operation for establishing links in call control unit 5 shown in Fig. 1 is next explained in more detail.

Fig. 2 shows an example of the composition of an RTCP sender report packet that is transmitted from LAN 10 at the time a link is established in call control unit 5 shown in Fig. 1.

As shown in Fig. 2, this composition is made up of a header portion and an information portion, and part of the information portion includes packet loss rate 14 and accumulated loss packet number 15, and this information is used for judging the start of congestion and the end of congestion in LAN 10.

Packet loss rate 14 in the RTCP sender report packet is a variable and is set to match a user's level of use.

When the establishment of a link to the partner node by way of LAN 10 is notified from call control unit 5, an

RTCP sender report packet is transmitted from LAN 10 to traffic control unit 7 by way of LAN driver 9 and RTP unit 8.

Packet loss rate 14 in the RTCP sender report
5 packet that is transmitted from LAN 10 by way of LAN driver 9 and RTP unit 8 is checked in traffic control unit 7. If packet loss rate 14 exceeds a predetermined set value, call control unit 5 is notified that packet loss rate 14 has exceeded the set value.

10 With each notification from traffic control unit 7 that packet loss rate 14 has exceeded the set value, the count value in a counter incorporated in call control unit 5 is incremented as the number of instances of congestion.

15 When packet loss rate 14 continuously exceeds the set value, the number of instances of congestion counted by the counter provided inside call control unit 5 may exceed a prescribed number, in which case call control unit 5 judges that LAN 10 is in a state of congestion,
20 and call control unit 5 notifies central processing unit 3 that LAN 10 is in a state of congestion.

Upon being notified from call control unit 5 that LAN 10 is in a congested state, central processing unit 3 notifies alternate routing control unit 4 that signal
25 path A, which is the first signal path, is in a congested state, following which inquiry is made regarding the

second signal path.

When inquiry is made from central processing unit 3 regarding the second signal path, alternate routing control unit 4 notifies central processing unit 3 that
5 the signal path that is set as the second signal path is signal path B, which is connected to the partner node by way of leased line network 13, which is in turn path-connected to leased line trunk 12.

Central processing unit 3 next directs switch
10 control unit 2 to reroute PCM voice signals that are transmitted from extension telephone 1, following which extension telephone 1 and leased line trunk 12 are path-connected by switch control unit 2, and PCM voice signals are subsequently transmitted from extension telephone 1
15 to the partner node by way of leased line network 13. Explanation is next presented regarding a system that reroutes from a signal path that passes by way of an IP network to another signal path in a case in which the IP network encounters a congested state while PCM voice
20 signals are being transmitted from extension telephone 1 to a partner node by way of the IP network.

Fig. 3 shows another embodiment of the IP network alternate routing system of the present invention.

In contrast to the system shown in Fig. 1, this
25 embodiment is provided with: announcement trunk 16 for notifying extension telephone 1 via switch control unit 2

that the transmission path has been switched; and signal path C, which is a signal path distinct from signal path B that connects extension telephone 1 and leased line trunk 12 by way of switch control unit 2.

5 The signal transmission operation in an IP network alternate routing system that is configured according to the foregoing explanation is next explained.

10 If it is judged in call control unit 5 that LAN 10 is in a congested state during the transmission of PCM voice signals from extension telephone 1 to a partner node via LAN 10, the congested state of LAN 10 is notified from call control unit 5 to central processing unit 3.

15 Upon receiving notification from call control unit 5 that LAN 10 is in a congested state, central processing unit 3 sends an inquiry to alternate routing control unit 4 regarding the second signal path, following which alternate routing control unit 4 notifies central processing unit 3 that signal path B is the second signal path.
20

 Central processing unit 3 instructs switch control unit 2 to reserve acquisition of leased line trunk 12 by way of signal path B, and at the same time, the link to the partner node via signal path C, leased line trunk 12, and leased line network 13 is established. Upon
25 establishing the link to the partner node via signal path

C, leased line trunk 12, and leased line network 13, a path between extension telephone 1 and announcement trunk 16 is connected by switch control unit 2, and the user that is using extension telephone 1 is notified from
5 announcement trunk 16 of the switch to signal path C.

The user that is using extension telephone 1 then manually switches switch control unit 2 to connect the path between extension telephone 1 and signal path C, and PCM voice signals are subsequently transmitted from
10 extension telephone 1 to the partner node by way of leased line trunk 12 and leased line network 13.

In the present invention as described in the foregoing explanation, when voice signals transmitted from the extension telephone are transmitted on an
15 Internet protocol network, the call control unit judges the state of congestion of the Internet protocol network based on a sender packet that is transmitted from the Internet protocol network, and if it judges that the Internet protocol network is in a congested state, a
20 switch control unit switches the signal path the voice signals such that the voice signals are rerouted from the Internet protocol network to a network other than the Internet protocol network, whereby the level of service of the system can be maintained.

25 Even in a case in which a problem occurs in the interface with the Internet protocol network in the

partner node, a link to the partner node is not established, and as a result, voice signals transmitted from the extension telephone are similarly automatically rerouted to a network other than the Internet protocol
5 network.

The packet loss rate for determining the state of congestion of an Internet protocol network is a variable and thus can be set to match a user's level of use.

Finally, voice signals that are transmitted from a
10 user that cannot tolerate deterioration in voice quality (for example, a VIP) can be automatically or manually rerouted to a public network or a TDM leased line, which is a circuit that is distinct from an Internet protocol network.

15 While preferred embodiments of the present invention have been described using specific terms, such description is for illustrative purposes only, and it is to be understood that changes and variations may be made without departing from the spirit or scope of the
20 following claims.

What is claimed is:

1. An Internet protocol network alternate routing system comprising an extension telephone, a plurality of networks provided with one Internet protocol network for transmitting voice signals from said extension telephone,
5 and an exchange for controlling connections between said extension telephone and said plurality of networks;

wherein said exchange, upon detecting a state of congestion of said Internet protocol network, automatically switches a network that is connected with
10 said extension telephone to a network other than said Internet protocol network.

2. An Internet protocol network alternate routing system according to claim 1, wherein said exchange comprises:

a plurality of signal paths for connecting
5 said plurality of networks with said extension telephone;
an alternate routing control unit for determining a network that is to be connected with said extension telephone;

a call control unit for establishing a link
10 between a network that has been determined by said alternate routing control unit and said extension telephone;

a traffic control unit for detecting a state

of congestion of a network for which a link has been
15 established by said call control unit; and

a switch control unit for controlling a
connection between said extension telephone and a signal
path among said plurality of signal paths that is
connected to a network, based on detection results in
20 said traffic control unit.

3. An Internet protocol network alternate routing
system according to claim 2, wherein said exchange
comprises:

a voice converter that, when a link has been
5 established with a partner node by said call control unit
and voice signals are transmitted from said extension
telephone by way of said switch control unit, converts
the transmitted voice signals to packets, assigns port
numbers to each of the packet voice signals, and outputs
10 the voice signals;

wherein said traffic control unit adds
Internet protocol addresses that are outputted from said
call control unit to voice signals outputted from said
voice converter to generate and output Internet protocol
15 packets;

and said exchange further comprises:

an RTP (Real-time Transport Protocol) unit for
adding RTP headers to Internet protocol packets outputted

from said traffic control unit and outputting the
20 Internet protocol packets; and

a network driver for transmitting on said
Internet protocol network Internet protocol packets to
which RTP headers have been added and that have been
outputted from said RTP unit.

4. An Internet protocol network alternate routing
system according to claim 3, wherein:

said RTP unit, in a case in which Internet
protocol packets to which RTP headers have been added are
5 transmitted from said Internet protocol network, removes
the RTP headers and outputs Internet protocol packets to
said traffic control unit;

said traffic control unit converts Internet
protocol addresses contained in Internet protocol packets
10 that are outputted from said RTP unit to port numbers of
ports in said voice converter, and outputs each of
Internet protocol packets that are assigned by port in
said voice converter to a respective port in said voice
converter; and

15 said voice converter converts Internet
protocol packets that are outputted from said traffic
control unit to voice signals, reconfigures converted
voice signals, and transmits to said extension telephone
by way of said switch control unit.

5. An Internet protocol network alternate routing system according to claim 2 wherein:

said traffic control unit detects a packet loss rate that is contained in a sender report packet
5 that is transmitted from said Internet protocol network and notifies said call control unit that said packet loss rate has exceeded a predetermined set value if said packet loss rate exceeds the set value; and

said call control unit is provided with a
10 counter in which a count value is incremented with each notification from said traffic control unit that said packet loss rate has exceeded a set value, and [said call control unit] does not establish a link between said Internet protocol network and said extension telephone if
15 the count value exceeds a predetermined set value.

6. An Internet protocol network alternate routing system according to claim 3 wherein:

said traffic control unit detects a packet loss rate that is contained in a sender report packet
5 that is transmitted from said Internet protocol network and notifies said call control unit that said packet loss rate has exceeded a predetermined set value if said packet loss rate exceeds the set value; and

said call control unit is provided with a

10 counter in which a count value is incremented with each
notification from said traffic control unit that said
packet loss rate has exceeded a set value, and [said call
control unit] does not establish a link between said
Internet protocol network and said extension telephone if
15 the count value exceeds a predetermined set value.

7. An Internet protocol network alternate routing
system according to claim 4 wherein:

said traffic control unit detects a packet
loss rate that is contained in a sender report packet
5 that is transmitted from said Internet protocol network
and notifies said call control unit that said packet loss
rate has exceeded a predetermined set value if said
packet loss rate exceeds the set value; and

said call control unit is provided with a
10 counter in which a count value is incremented with each
notification from said traffic control unit that said
packet loss rate has exceeded a set value, and [said call
control unit] does not establish a link between said
Internet protocol network and said extension telephone if
15 the count value exceeds a predetermined set value.

8. An Internet protocol network alternate routing
system according to claim 5 wherein the packet loss rate
in said sender report packet is variable.

9. An Internet protocol network alternate routing system according to claim 6 wherein the packet loss rate in said sender report packet is variable.

10. An Internet protocol network alternate routing system according to claim 7 wherein the packet loss rate in said sender report packet is variable.

11. An Internet protocol network alternate routing system according to claim 2 wherein said switch control unit can be manually switched.

12. An Internet protocol network alternate routing system according to claim 3 wherein said switch control unit can be manually switched.

13. An Internet protocol network alternate routing system according to claim 4 wherein said switch control unit can be manually switched.

14. An Internet protocol network alternate routing system according to claim 5 wherein said switch control unit can be manually switched.

15. An Internet protocol network alternate routing

system according to claim 6 wherein said switch control unit can be manually switched.

16. An Internet protocol network alternate routing system according to claim 7 wherein said switch control unit can be manually switched.

17. An Internet protocol network alternate routing system according to claim 8 wherein said switch control unit can be manually switched.

18. An Internet protocol network alternate routing system according to claim 9 wherein said switch control unit can be manually switched.

19. An Internet protocol network alternate routing system according to claim 10 wherein said switch control unit can be manually switched.

20. An Internet protocol network alternate routing system according to claim 11 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

21. An Internet protocol network alternate routing system according to claim 12 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

22. An Internet protocol network alternate routing system according to claim 13 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

23. An Internet protocol network alternate routing system according to claim 14 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

24. An Internet protocol network alternate routing system according to claim 15 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a

congested state.

25. An Internet protocol network alternate routing system according to claim 16 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

26. An Internet protocol network alternate routing system according to claim 17 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

27. An Internet protocol network alternate routing system according to claim 18 further comprising an announcement trunk for reporting switching of said signal path to said extension telephone when a network
5 determined by said alternate routing control unit is in a congested state.

28. An Internet protocol network alternate routing system according to claim 19 further comprising an announcement trunk for reporting switching of said signal

path to said extension telephone when a network
5 determined by said alternate routing control unit is in a
congested state.

ABSTRACT OF THE DISCLOSURE

An Internet protocol network alternate routing system comprises: a traffic control unit for detecting the state of congestion of an LAN, which is an Internet
5 protocol network; a call control unit for establishing a link between an extension telephone and the LAN based on the detection results of the traffic control unit; a central processing unit for designating the signal path of voice signals that are transmitted from the extension
10 telephone based on the results of establishing a link in the call control unit; and a switch control unit for switching the signal path that connects a path to the extension telephone. In a case in which the call control unit does not establish a link, voice signals that are
15 transmitted from the extension telephone are rerouted by switching performed by the switch control unit to a leased line network by way of a leased line trunk.

Fig. 1

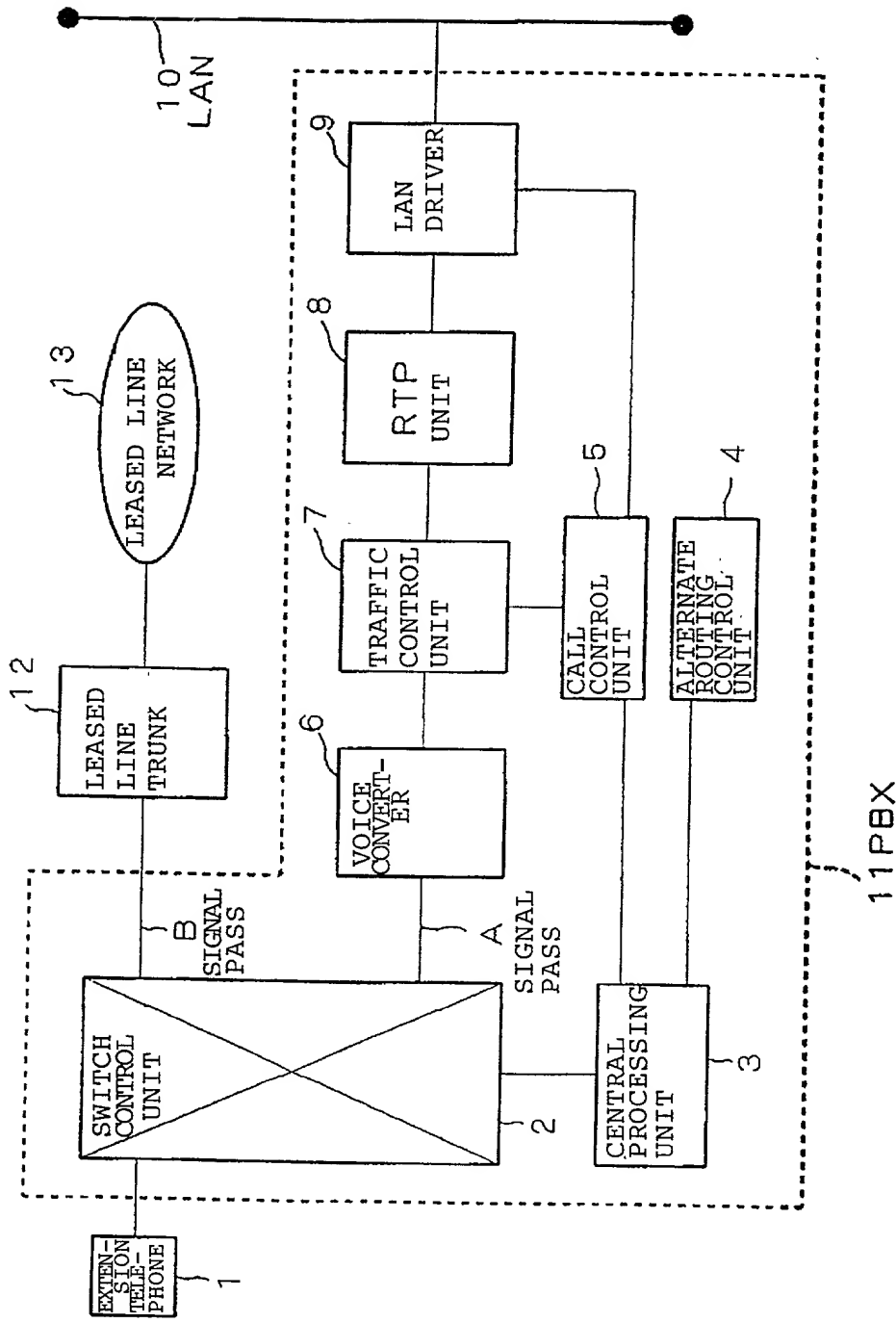


Fig. 2

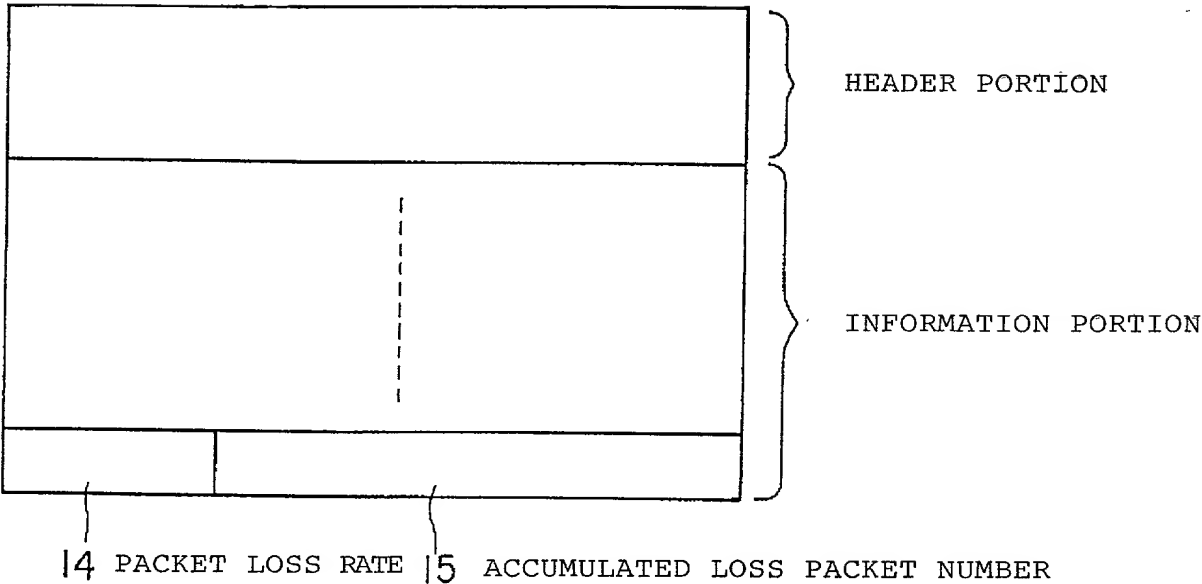
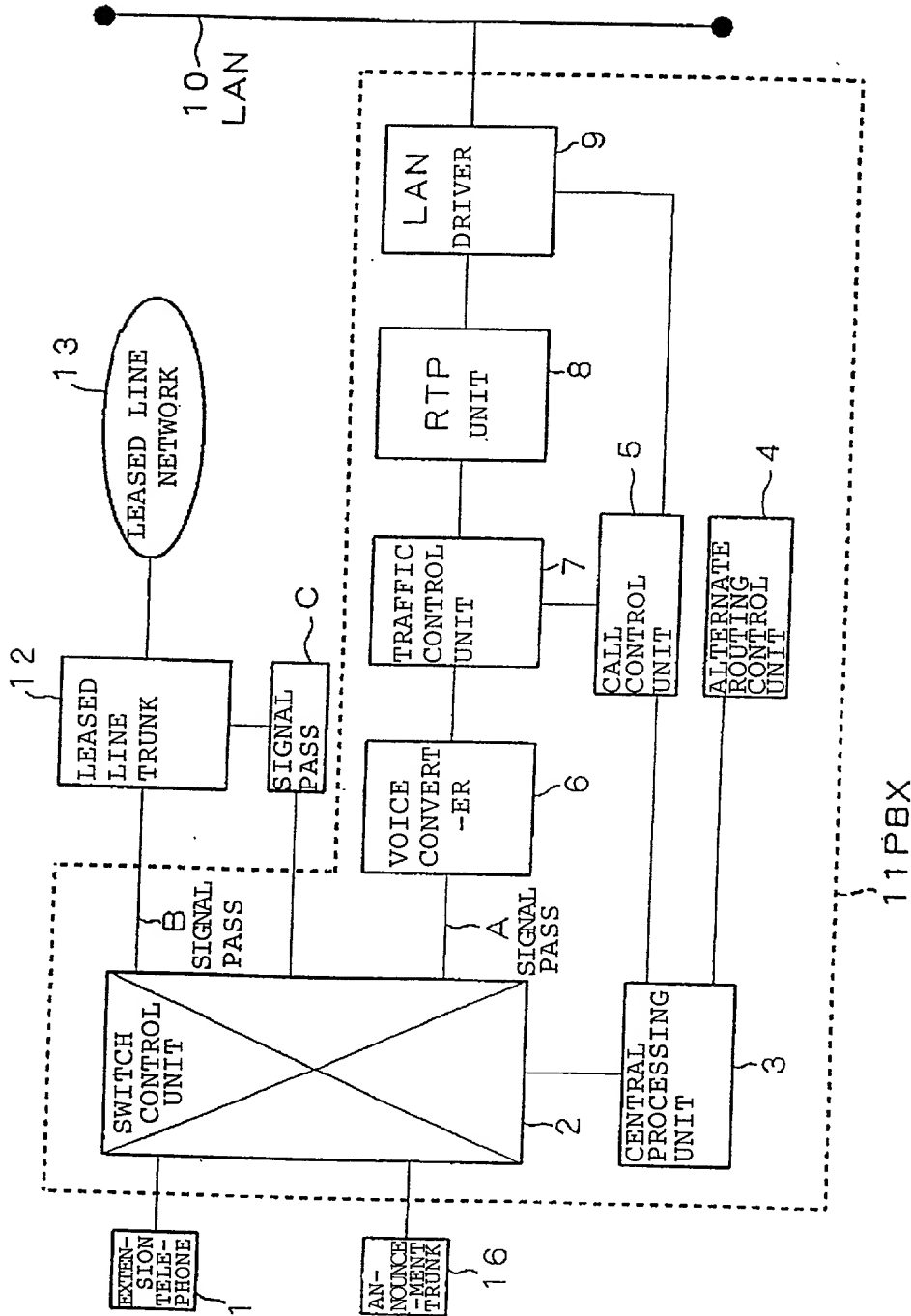


Fig. 3



DECLARATION AND POWER OF ATTORNEY

As a below named inventor, I hereby declare that:

My residence, post office address and citizenship are as stated below next to my name.

I believe I am the original, first and sole inventor (if only one name is listed below) or an original, first and joint inventor (if plural names are listed below) of the subject matter which is claimed and for which a patent is sought on the invention entitled:

Internet Protocol Network. Alternate Routing System

the specification of which is attached hereto unless the following box is checked:

☐ was filed on _____ as United States Application Number or PCT International Application Number _____ and was amended on _____ (if applicable).

I hereby state that I have reviewed and understand the contents of the above-identified specification, including the claims, as amended by any amendment referred to above.

I acknowledge the duty to disclose information which is known by me to be material to patentability as defined in Title 37, Code of Federal Regulations § 1.56.

I hereby claim foreign priority benefits under Title 35, United States Code, § 119(a)-(d) or § 365(b) of any foreign application(s) for patent or inventor's certificate, or § 365(a) of any PCT International application which designated at least one country other than the United States, listed below and have also identified below any foreign application for patent or inventor's certificate, or PCT International application having a filing date before that of the application on which priority is claimed:

PRIOR FOREIGN APPLICATION(S)

NUMBER	COUNTRY	DAY/MONTH/YEAR FILED	PRIORITY CLAIMED
11-154078	Japan	01/06/1999	Yes

I hereby claim the benefit under Title 35, United States Code § 119(e) of any United States provisional application(s) listed below.

APPLICATION NO.	FILING DATE

I hereby claim the benefit under Title 35, United States Code, § 120 of any United States application(s), or § 365(c) of any PCT International application designating the United States, listed below and, insofar as the subject matter of each of the claims of this application is not disclosed in the prior United States or PCT International application in the manner provided by the first paragraph of Title 35, United States Code, § 112. I acknowledge the duty to disclose information which is known by me to be material to patentability as defined in Title 37, Code of Federal Regulations § 1.56 which became available between the filing date of the prior application and the national or PCT International filing date of this application:

APPLICATION SERIAL NO.	FILING DATE	STATUS: PATENTED, PENDING, ABANDONED

I hereby appoint as my attorneys, with full powers of substitution and revocation, to prosecute this application and transact all business in the Patent and Trademark Office connected therewith: Stephen A. Bent, Reg. No. 29,768; David A. Blumenthal, Reg. No. 26,257; John J. Feldhaus, Reg. No. 28,822; Donald D. Jeffery, Reg. No. 19,980; Eugene M. Lee, Reg. No. 32,039; Peter G. Mack, Reg. No. 26,001; Brian J. McNamara, Reg. No. 32,789; Sybil Meloy, Reg. No. 22,749; George E. Quillin, Reg. No. 32,792; Colin G. Sandercock, Reg. No. 31,298; Bernhard D. Saxe, Reg. No. 28,665; Charles F. Schill, Reg. No. 27,590; Richard L. Schwaab, Reg. No. 25,479; Arthur Schwartz, Reg. No. 22,115; Harold C. Wegner, Reg. No. 25,258.

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I hereby declare that all statements made herein of my own knowledge are true and that all statements made on information and belief are believed to be true; and further that these statements were made with the knowledge that willful false statements and the like so made are punishable by fine or imprisonment, or both, under Section 1001 of Title 18 of the United States Code and that such willful false statements may jeopardize the validity of the application or any patent issued thereon.

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